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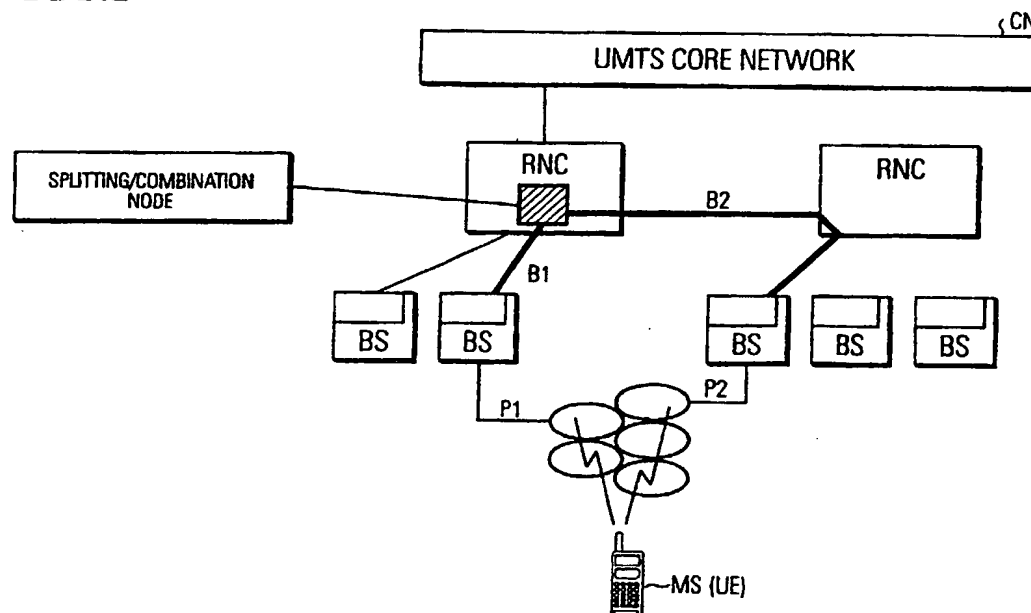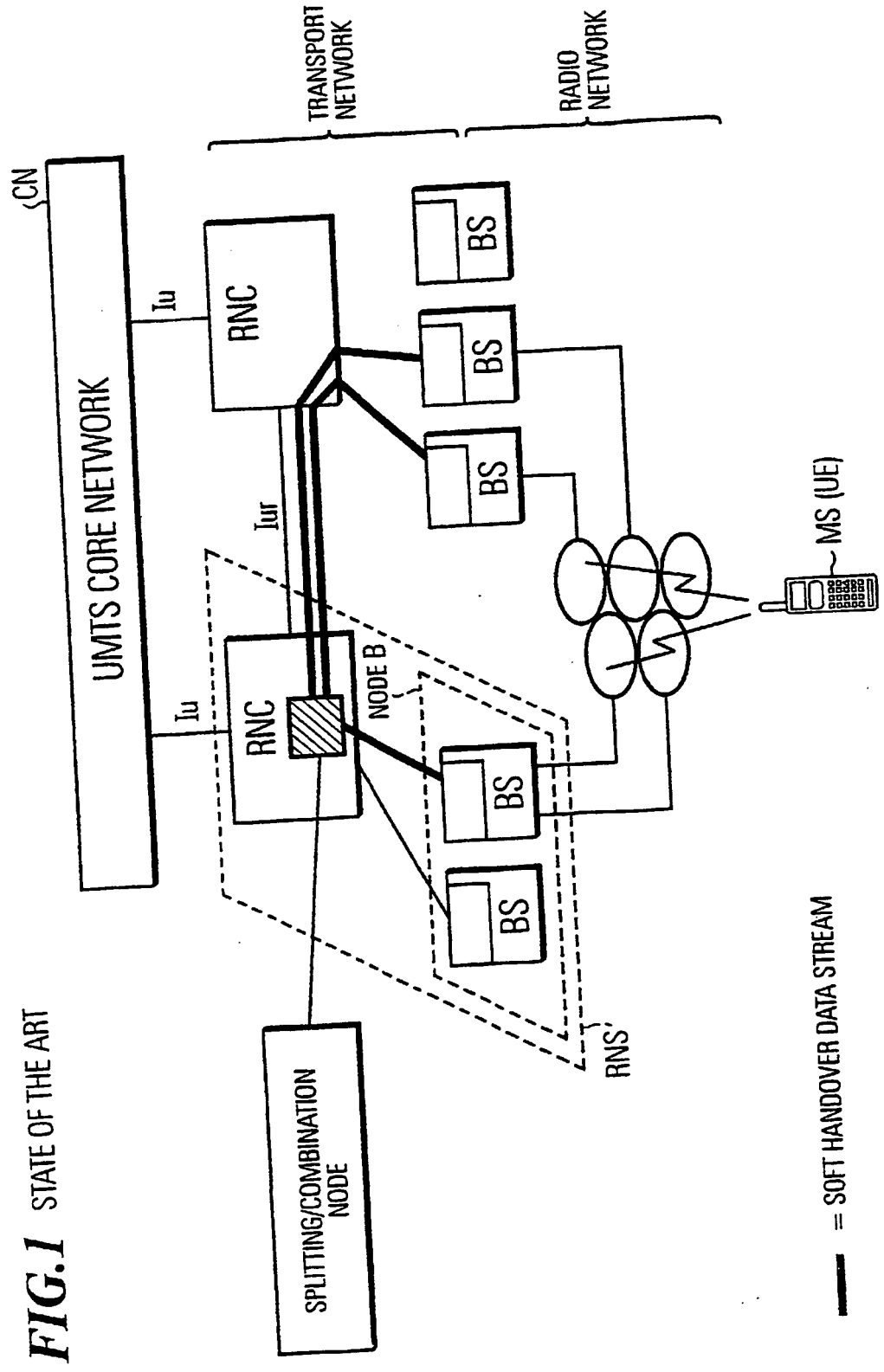Networking Conference, 1999, vol.2, p781-786.

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(54) Abstract Title
Prioritising packets in a mobile phone network soft handover method

(57) In a mobile access network using packet transmission such as a UMTS (Universal Mobile Telephone System), a mobile station MS has a plurality of active soft handover connections (macrodiversity) to the network, each having a separate path from the mobile to a splitting/recombination node (RNC). Each path consists of a radio channel to a base station and a transport channel from the base station to the RNC. The quality of each radio channel is measured at the base station and a higher priority is given to connections having higher quality radio channels. Packets are marked according to the priority assigned to the channel via which they travel. Buffer management is effected in all nodes through which the packets travel to ensure that a lower packet loss ratio is experienced by higher priority packets.
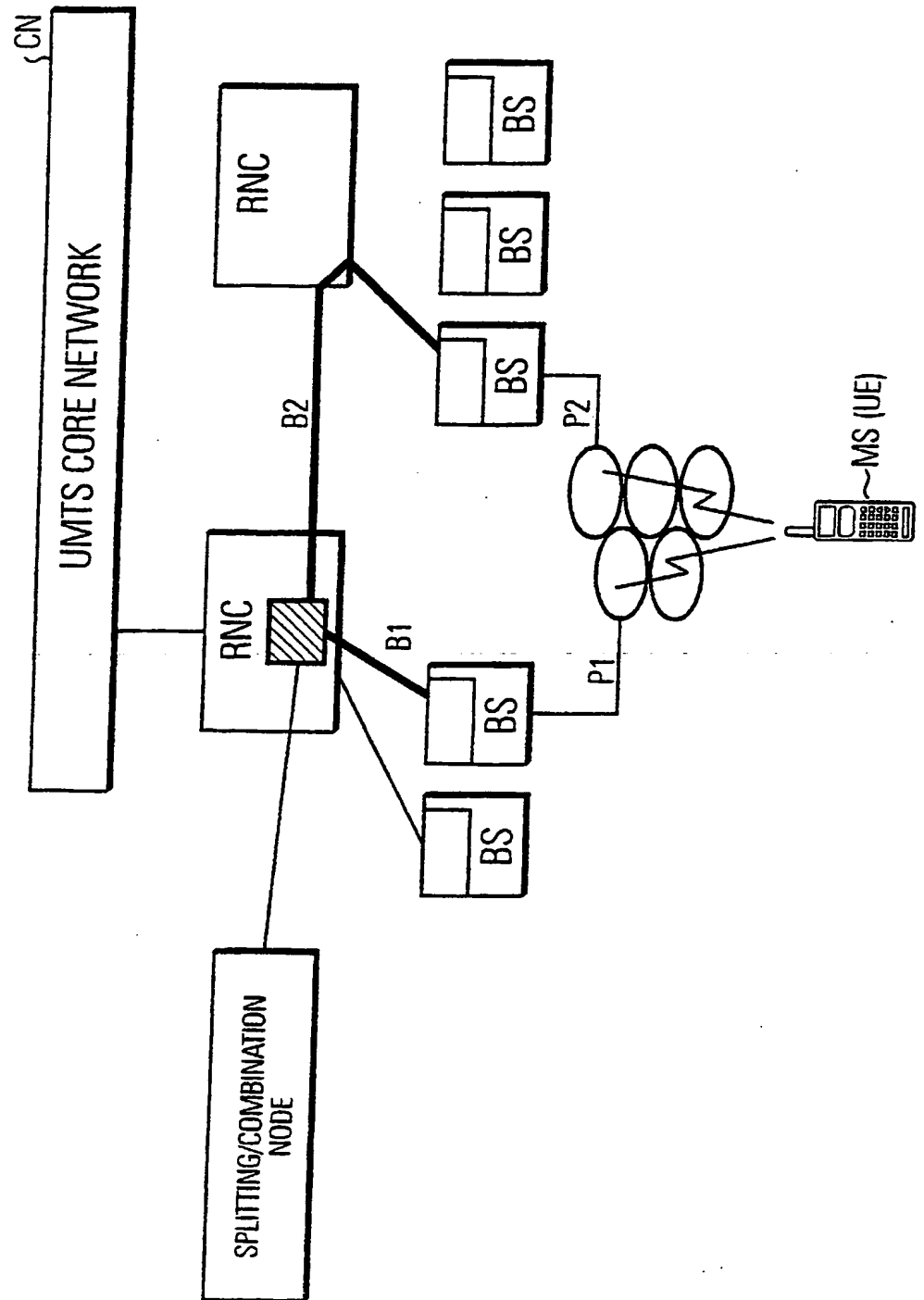
FIG.2

FIG.1 STATE OF THE ART

UMTS CORE NETWORK

CN

TRANSPORT NETWORK

RADIO NETWORK

RNC

Iu

BS

BS

BS

Iur

RNC

Iu

NODE B

BS

BS

RNS

MS (UE)

SPLITTING/COMBINATION NODE

= SOFT HANDOVER DATA STREAM

FIG.2

FIG.3

FIG.4

PACKET IN

CURRENT OCCUPANCY
LEVEL 0

FIFO BUFFER

O1

O2

O3

MAXIMUM
OCCUPANCY
LEVELS

FIG.5

FIG.6

TOTAL ERROR RATIO RECOMBINATION (E)

TRANSPORT LOSS RATIO ON FIRST SHO CONNECTION (B1)

CONNECTION WITH BETTER RADIO CHANNEL QUALITY IS PRIORITISED

NO DIFFERENTIATION AMONG TRANSPORT CONNECTIONS

FIG.7

RELATIVE GAIN OF SHO TRANSPORT PRIORITISATION (G)

TRANSPORT PACKET LOSS RATIO (B)
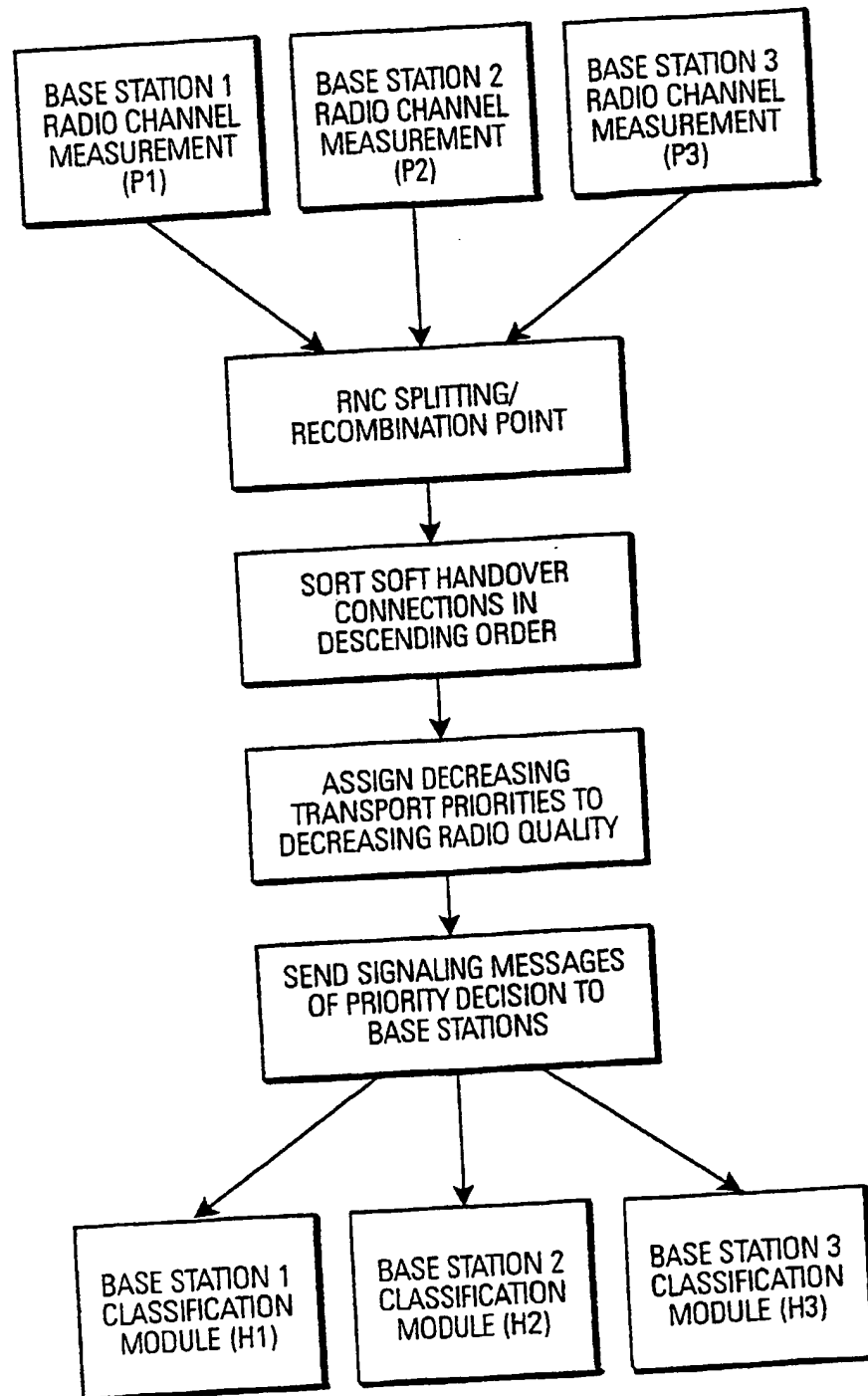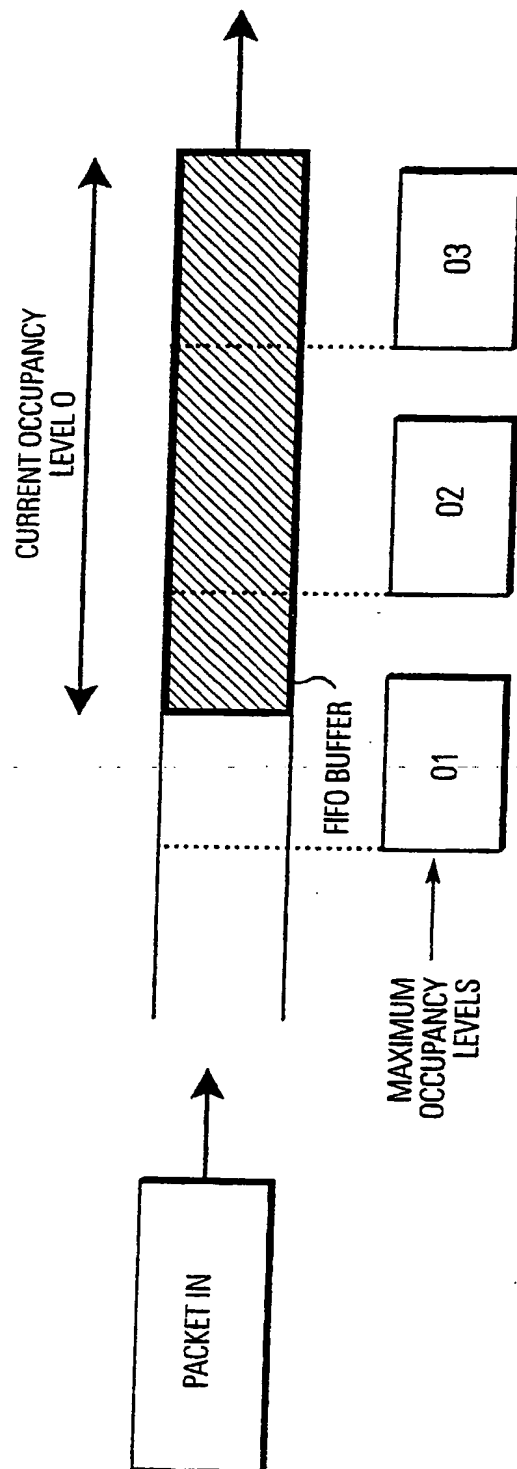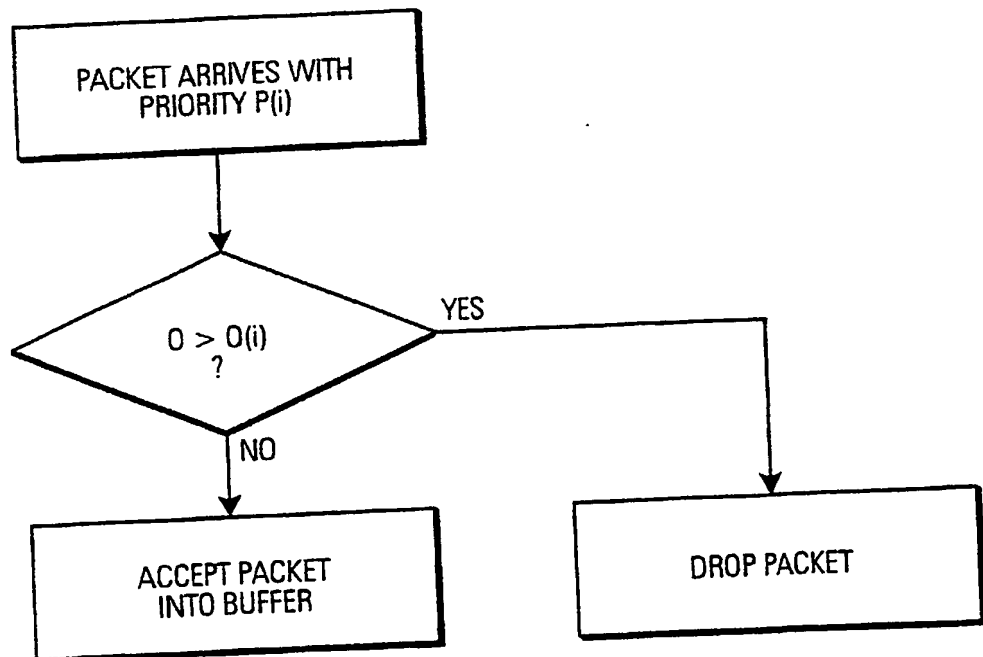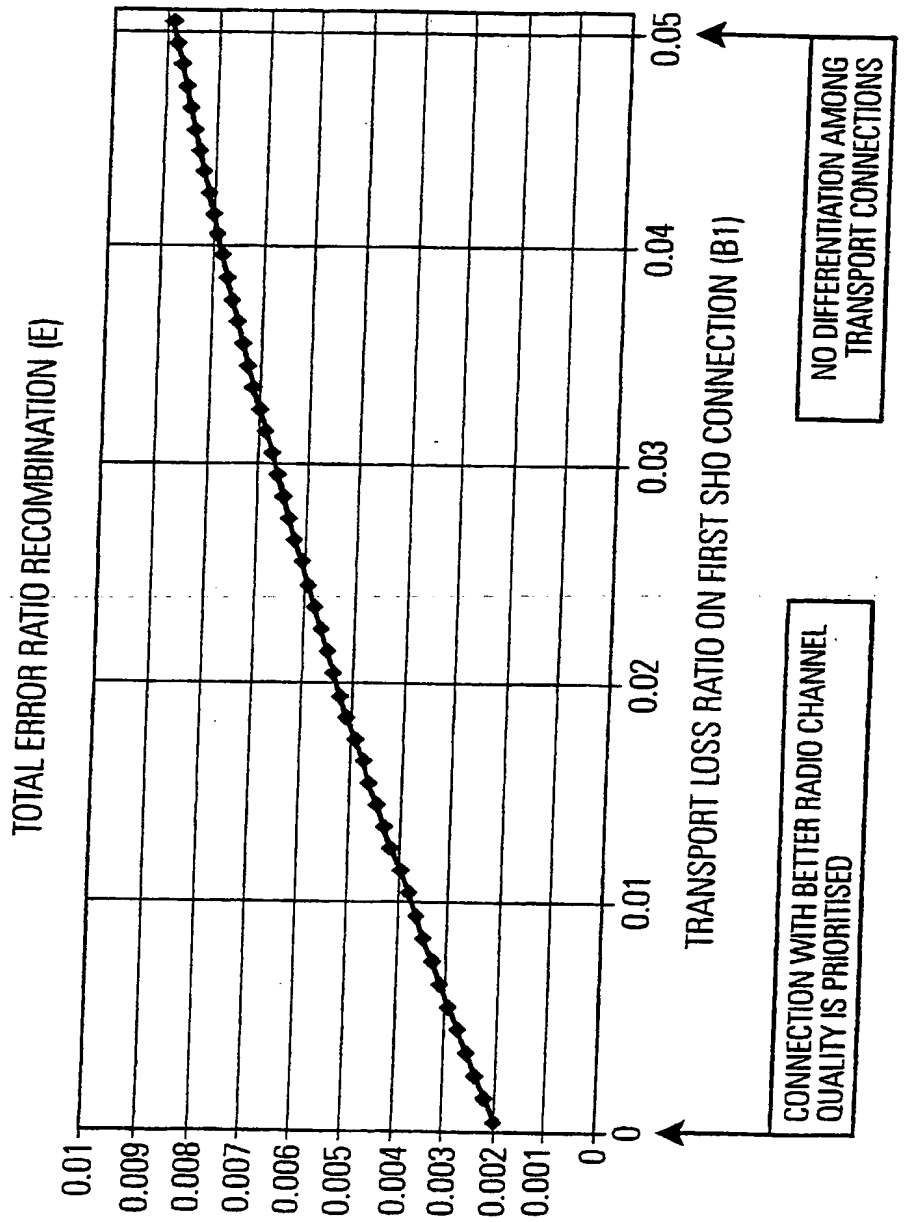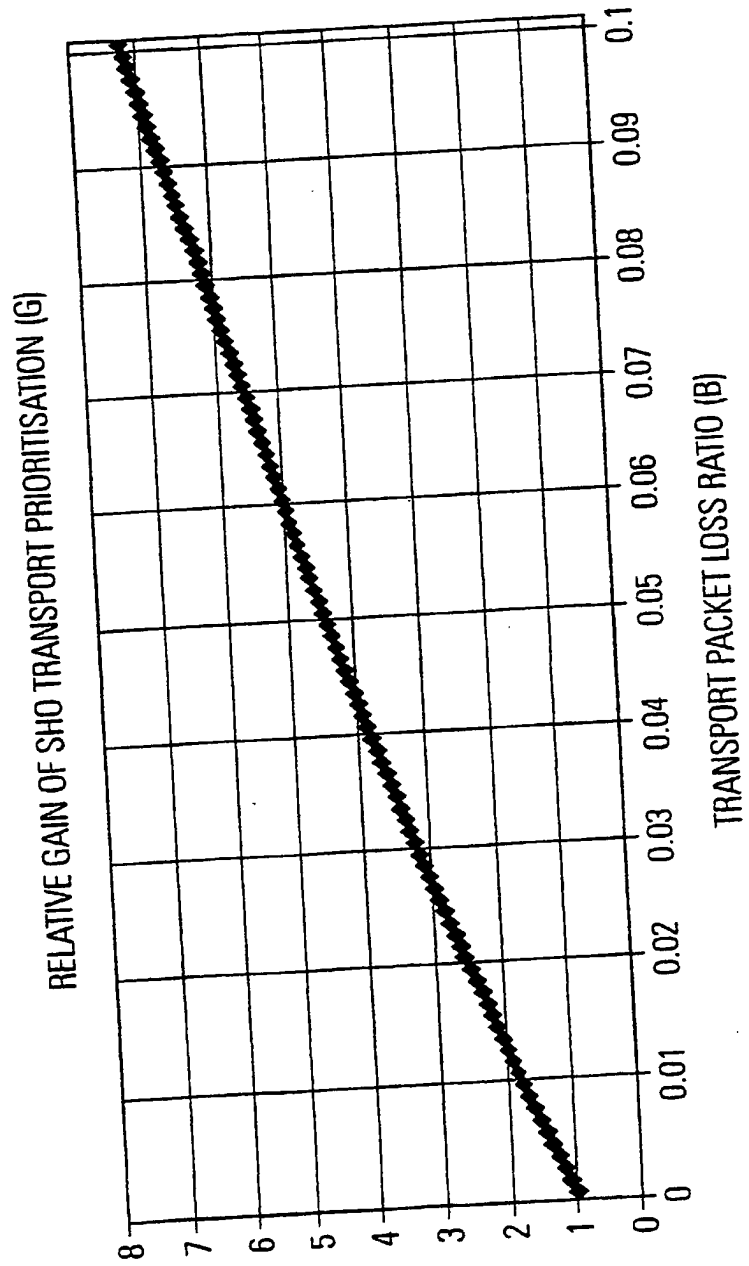
Short Title

Mobile access networks

5   Technical field of the invention

The present invention relates to mobile access networks and is concerned in particular with soft handover in such networks.

10

Technical background and related art

A third generation mobile communications system, such as a UMTS (Universal Mobile Telephone System) system, can be
15   considered conceptually as comprising three major components or parts, as illustrated in Figure 1. These are a core network (CN), a transport or radio access network and a radio network.

20   The core network CN is the fixed infrastructure of the system which comprises entities to support mobility management, database operations and internetworking with other networks, such as PSTN networks (Public Switched Telephone Networks)..
25

The transport or radio access network comprises entities that provide functionality to support radio coverage, in particular comprising radio interface support, radio resource management and radio operation and maintenance
30   functions.   In a UMTS system, the transport network or UTRAN (UMTS terrestrial radio access network) comprises a set of radio network subsystems (RNS's) each having a Radio Network Controller (RNC) and one or more Node B's (embodied as base stations (BS) for example) - as indicated by broken
35   lines on the left of Figure 1.   Each RNS of the UTRAN is connected to the core network (CN) through a logical interface (Iu), and different RNS's may connected one to

another through a further logical interface (Iur).

The radio network is the radio channels between the Node B's (base stations BS) and mobile stations (MS) or user equipment (UE).

In third generation mobile access networks, such as for example UMTS networks, use is made of the macrodiversity (macroscopic diversity) technique, which enables a mobile station or user equipment (MS or UE) to be continuously connected to (communicating with) two or more base stations (BS) via a set of independent radio paths, as schematically illustrated in Figure 1. By combining received signals from the two or more independent radio paths of the set at a splitting/recombination node (in a radio network controller node, RNC, of the transport or radio access network) significantly better service quality can be achieved.

In this respect it will be understood that an important feature of third generation networks is that all data are sent in data packets (for example TCP/IP packets or ATM "cells") which "packet" method enables the utilisation of statistical multiplexing gain (when using the macrodiversity technique). This "packet" method also enables the admission of more connections, as compared to traditional circuit switched networks, but at the price that a certain ratio of packets may be lost due to traffic rate fluctuations.

Soft handover is a handover technique related to macrodiversity, in which handover is executed by adding a new connection to the macrodiversity set of radio path connections to base stations (Node B's), and releasing an old connection.

Thus, when macrodiversity is employed (macrodiversity may also be referred to simply as soft handover or soft

handover mode) one logical connection uses more than one physical connection between the mobile station (MS) and the splitting/recombination node. These physical connections are called soft handover connections of a given logical

5   connection. As schematically illustrated in Figure 1, soft handover connections may traverse several nodes in the transport network between the base stations and the splitting/recombination node.

10   Depending on the technology used in the transport network, different implementations are used to support soft handover functionality in the network, e.g. with ATM technology AAL2 (ATM Adaptation Layer 2) connections may be used, with MPLS (Multiprotocol Label Switching) technology label paths may

15   be used, and with TCP/IP networks IntServ (Integrated Services in the Internet) or DiffServ (Differentiated Services in the Internet) architectures may be used.

However, regardless of the technology used, in the state of

20   the art, when a mobile station (MS) is entering a soft handover state in a third generation mobile access network, an appropriate transport network connection must be established between the base station (BS) (or Node B) and the soft handover splitting/recombination node. Since the

25   air interface is the capacity limiting part of the access network in third generation access networks, the soft handover connection must guarantee very low packet loss ratio in the transport network. This requires very rigid admission control strategy in the transport nodes.

30

Further, in the state of the art, the same amount of transport network resources and the same quality of service guarantees are required for all soft handover connections regardless of the connections` radio quality and the number

35   of soft handover connections existing for the service.

Summary

Problems addressed by the present invention

5   As a consequence of the very rigid admission control strategy in the transport nodes, for soft handover connections in the state of the art, the larger part of possible statistical multiplexing gain is lost due to the necessary overdimensioning of the transport links called

10   for by this rigid admission control.

Solution provided by the present invention

The present invention provides a method of soft handover in

15   mobile access network using packet transmission, in which a mobile host has a plurality of active soft handover connections (macrodiversity) to the access network, those connections having respective independent radio paths from the mobile host to a splitting/recombination node of the

20   mobile access network, each radio path having a radio channel to/from the mobile host to the node of the mobile access network which is in radio signal communication with the mobile host, and a transport channel to/from that radio signal communication node to the splitting/recombination

25   node (RNC). In the method the quality of the radio channel is measured at the radio communication signal node concerned, for each of the soft handover radio paths; a transport priority decision is made in the splitting/recombination node, the decision giving higher

30   priorities to the soft handover connections having higher quality radio channels; packet classification is effected at the or each sending node, a sending node being a radio signal communication node in the uplink direction, and the sending node being the splitting/recombination node in the

35   downlink direction, each packet to be sent being marked according to the assigned priority level of the radio channel via which it has travelled or is to travel; and

buffer management is effected in all nodes handling the marked packets, in accordance with the assigned priority levels of the packets, to ensure that a lower packet loss ratio is experienced by higher priority packets.

5   The present invention also provides a soft handover system.

Advantages of the invention

10   The present invention has many advantages. For example, with the present invention, by monitoring or controlling the transmission loss ratios of different soft handover connections, the error ratio after recombination can be significantly improved even if the transmission network is
15   congested. Further, for the same transport network design a higher level of network load can be maintained with soft degradation of service quality with increasing traffic demand unlike in the prior art, where there is a rigid requirement on transport quality. The present invention
20   thus enables more soft handover connections to be maintained without rigid extra requirements on the transport, and better radio channel efficiency can be achieved with lower radio interference levels. As a consequence the quality and capacity of the air interface
25   can be influenced, providing statistically more soft-handover connections available for combining.

Brief description of the drawings

30   Figure 1   is a schematic illustration of a third generation UMTS network;

       Figure 2   is a schematic illustration for assistance in explanation of an embodiment of the present
35   invention;

       Figure 3   is a flow chart illustrating procedures carried

out in an embodiment of the present invention;

Figure 4 schematically illustrates a buffer management procedure as may be applied in an embodiment of the present invention;

Figure 5 schematically illustrates a further aspect of the buffer management procedure or Figure 4;

Figure 6 is a graph illustrating beneficial effects of an embodiment of the present invention; and

Figure 7 is a graph further illustrating beneficial effects of an embodiment of the present invention.

The soft handover recombination efficiency in case of lossy transmission paths can be estimated. The calculations are applicable for both uplink and downlink recombination.

For simplicity it is assumed, as illustrated in Figure 2, that there are two soft handover connections active for a mobile host. The error ratios of the two radio channels concerned are denoted by P1 and P2 in Figure 2. Furthermore, the packet loss ratios on the two transmission paths up to the recombination node are denoted by B1 and B2 in Figure 2.

At the recombination point the observed error ratio is thus increased for both soft handover connections, since a certain ratio of packets carrying useful information is lost in the transport network. The path error ratio for both the radio and the transport network parts is thus approximately (for small ratios):

(1a) $E1 = P1 + B1$

(1b) E2 = P2 + B2

The recombination is successful if correct information arrives from either connection, i.e. the error ratio for

5   the information received from the mobile station after recombination can be approximated by: -

(2a) E = E1 * E2

10   or

(2b) E = (P1 + B1) * (P2 + B2)

In an access network in accordance with the state of the

15   art, the relation of B1 and B2 is not controlled or checked: only a very rigid admission procedure or packet scheduling method, as mentioned above, is applied to limit these values, with no differentiation on the basis of (i.e. taking no account of) the soft handover radio path

20   properties, i.e. P1 and P2.

Further, in the state of the art, packets on the transport network are not differentiated on the basis of the soft handover connection properties. This means that all packets

25   of a certain service (e.g. voice) are treated the same way in the buffers along the connections` paths. Consequently the packet loss probability is equal for all soft handover connections (for a given service).

30   However, the inventors of the present invention have had the insight that (see the above formula (2b) that the final loss ratio after recombination is in direct relation of all four (P1, P2, B1, B2) values. They have had the further insight that network performance can be optimised by

35   optimally controlling or monitoring these values.

Thus, in the state of the art, macrodiversity or soft

handover in a (logical) radio channel can be used to significantly improve the radio quality by statistically combining independent radio connections. In accordance with the present invention, it might be said, a similar

5 principle is applied in respect of the transport network, with similar expected improvements.

Embodiments of the present invention are not generally concerned with soft handover between sectors of the same

10 base station (also known as softer handover). Softer handover requires no additional transport network connection, as splitting/recombination shall be performed in the base station. Thus, the description below will concentrate on soft handover, as opposed to softer

15 handover, which is the primary concern of embodiments of the present invention.

It is observed that embodiments of the present invention are applicable in both uplink (combining) and downlink

20 (splitting) directions.

The present invention is applicable for the soft handover connections in the transport network of a given logical connection between the mobile station and the

25 splitting/recombination node.

In accordance with the present invention, packets containing radio frames are marked for appropriate packet handling at the nodes which the packets traverse. In TCP/IP

30 networks the IP Type of Service (TOS) or DiffServ (DS) field can be used for differentiation in packet handling, in ATM based access networks the ATM cell CLP (Cell Loss Priority) bit can be used for this purpose. As radio path properties may change very fast, these settings may be

35 determined on a packet-by-packet basis depending on the measured path properties.

In accordance with the present invention, provisions are made for:-

- •Radio channel quality measurement at the base station
5
- •Transport priority decision in the RNC
- •Packet service classification at the sending node, in uplink this is the base station, in downlink this is the RNC node.
- •Buffer management method in all transmission nodes
10 handling packets according to priority

Radio channel quality measurement and Transport priority

In embodiments of the present invention, the measurement of
15 radio channel quality is performed in the base stations (BS) (or Node B's). This measurement does not have to be the direct error ratio, but any related measurement e.g. radio interference measurement can be used equivalently. The radio channel measurements are sent to the RNC (Radio
20 Network Controller) having the splitting/recombination node periodically.

In embodiments of the present invention, the RNC decides transport priority (H) based on the measurements: the best
25 radio channel of a mobile terminal always receives the highest priority; e.g. for P1 < P2 < P3 the transport priorities would be H1 > H2 > H3.

Packet classification
30

In embodiments of the present invention, packet classification is performed at the sending node. In an uplink this is the base station, in a downlink this is the RNC node. During classification, each packet to be sent is
35 marked according to the assigned priority level, e.g. priority H2 is set for packets sent on the P2 connection.

The flowchart of Figure 3 illustrates the steps involved in the an embodiment of the present invention in relation to radio channel quality measurement and transport priority and packet classification, in this case in the base stations.

Buffer management

In embodiments of the present invention, in every node of the transport network the packets are handled according to the set priorities. The buffer management ensures that a lower packet loss ratio is experienced by higher priority packets. It should preferably also be ensured that packets belonging to the same connection are not reordered even if they have different priorities, since it would cause confusion if a packet sent earlier arrives later at the destination node.

In a preferred embodiment of the present invention the following buffer management scheme is used, which meets the above requirements:

A single First In First Out buffer is used for all packets, all priorities and all connections of a given service type. Priority handling is ensured by setting different maximum occupancy levels (e.g., O1, O2, O3) for different priorities (e.g. H1, H2, H3). If H1 > H2 > H3, then O1 > O2 > O3 respectively.

If a packet arrives at the buffer it is dropped if the current occupancy level (O) is higher than the appropriate maximum occupancy level of the packet priority.

This buffer management scheme is schematically illustrated in Figures 4 and 5.

This buffer management scheme ensures that packets of

higher priority experience a lower packet loss ratio and that packets are not reordered.

In an embodiment of the present invention the transmission loss ratio of soft handover connections is controlled or monitored in an appropriate way such that the error ratio after recombination can be significantly improved even if the transmission network is highly congested. Alternatively, in an embodiment of the present invention, for the same transport network design a higher network load can be maintained with soft-degradation in service quality with increasing traffic demand, unlike prior art proposals where there is a rigid requirement on transport quality. Further, with the employment of an embodiment of the present invention, it is unnecessary or less necessary to release existing or block new soft handover requests due to high transport congestion, and thus better radio channel efficiency can be achieved with lower radio interference levels.

Below there is presented a simple but realistic network scenario.

It is well known that purely by buffer management or scheduling methods the gross packet loss ratio on the transport network cannot be influenced. The average loss ratio on the transport network is denoted by B. By means of buffer management, only the distribution of losses can be controlled, i.e. a preference for one connection or the other can be provided, but the average loss probability remains the same:

(3) $B = (B1 + B2)/2$

Substituting this result in the expression for the error ratio after recombination we get:

(4) $E = (P1 + B1) * (P2 + 2*B - B1)$

If one soft handover radio channel has lower radio channel error ratio, e.g. P1 < P2, then the optimum value of E is obtained if B1 = 0, i.e. the path which has better radio quality should be prioritised in the transport network. In this case

(5) $Epri = P1 * (P2 + 2*B)$

The gain with transport prioritisation is thus

(6a) $G = E/Epri$

(6b) $G = ((P1 + B1)*(P2 + B2))/(P1 * (P2 + 2*B))$

In prior art systems there is no differentiation between soft handover connections, and the packet loss ratio in the transport network is equal for all connections, i.e.

(7) $B1 = B2 = B$

Consequently the gain of provided by the present invention over the prior art is:-

(8) $G = ((P1 + B)*(P2 + B))/(P1 * (P2 + 2*B))$

To provide a concrete example, it is assumed that there are two soft handover connections for the mobile station. The error ratio of the better radio channel is P2 = 0.01. The second radio channel has a worse ratio of P2 = 0.1. The average transmission loss probability is B = 0.05.

P1 = 0.01, P2 = 0.1, B = 0.05

Calculating the error ratio after recombination for different levels of prioritisation between soft handover

connections results are obtained as illustrated in Figure 6, which shows total error ratio after recombination (E) on the vertical axis, plotted against the transport loss ratios on the first soft handover (SHO) connection B1,

5   corresponding to different prioritisations of the connection with the better radio channel quality, i.e. prioritisation of the better radio channel vis-à-vis no prioritisation (as per the prior art).

10 The results shows that the lowest error ratio after recombination can be achieved if the transport packet loss ratio of the better radio channel is minimised (B1 = 0), even if this causes the other soft handover connection to experience significantly worse transport service quality.

15 By keeping B1=0, E=0.002 can be achieved.

On the other hand, the prior art would achieve approximately E=0.01 which is significantly worse.

20 The relative gain (G) provided by the present invention in this example, in comparison with the prior art, as a function of the transport network overall congestion (transport packet loss ratio B) Figure 8.

25 Thus, in accordance with the present invention, by controlling or monitoring the transmission loss ratios of different soft handover connections, the error ratio after recombination can be significantly improved even if the transmission network is congested. Further, for the same

30 transport network design a higher level of network load can be maintained with soft degradation of service quality with increasing traffic demand unlike in the prior art, where there is a rigid requirement on transport quality.

35 The present invention enables more soft handover connections to be maintained without rigid extra requirements on the transport, thus better radio channel

efficiency can be achieved with lower radio interference levels.

As a consequence the present invention can influence the quality and capacity of the air interface, providing statistically more soft-handover connections available for combining.

It will be appreciated that the present invention requires the interworking of several network elements such as base station (Node B), routers/switches in the access network and the RNC. However, the major components are readily available in current platforms. For operation it is required that radio channel measurements are signalled to the RNC periodically. This is already effected in some prior art systems for different reasons, and can probably be reused for the present invention. In an embodiment of the present invention the end nodes of the connections have to be able to determine appropriate transport priority levels, which can be carried our by software. Packet marking of transport connections is supported by most current routers/switches. The buffer management method proposed above for use in an embodiment of the present invention may require new hardware design, but if the requirement on packet reordering is not so rigid, other buffer management methods may be alternatively applied, such as, for example, WFQ (Weighted Fair Queueing), RIO (Random Early Detection - RED - with In/Out Bit).

CLAIMS

1. A method of soft handover in a mobile access network using packet transmission, in which a mobile host has a
5 plurality of active soft handover connections (macrodiversity) to the access network, the soft handover connections having respective independent radio paths from the mobile host (MS) to a splitting/recombination node (RNC) of the mobile access network, each radio path having
10 a radio channel to/from the mobile host (MS) to the node (BS, Node B) of the mobile access network which is in radio signal communication with the mobile host (MS), and a transport channel to/from that radio signal communication node (BS, Node B) to the splitting/recombination node
15 (RNC), the method comprising:-
     measuring the quality of the radio channel at the radio communication signal node (BS, Node B) concerned, for each of the soft handover radio paths;
     making a transport priority decision in the
20   splitting/recombination node (RNC), the decision giving higher priorities to the soft handover connections having higher quality radio channels;
     effecting packet classification at the or each sending node, a sending node being a radio signal
25   communication node (BS, Node B) in the uplink direction, and the sending node being the splitting/recombination node (RNC) in the downlink direction, each packet to be sent being marked according to the assigned priority level of the radio
30   channel via which it has travelled or is to travel; and
     effecting buffer management in all nodes handling the marked packets, in accordance with the assigned priority levels of the packets, to ensure that a lower
35   packet loss ratio is experienced by higher priority packets.

2. A method as claimed in claim 1, wherein the buffer management ensures that packets belonging to the same soft handover connection are not reordered even if they have different priorities.

3. A method as claimed in claim 1 or 2, wherein the buffer management employs, in a sending node concerned, a single First-In First-Out buffer for all packets, all priorities and all connections of a given service type, priority handling of higher priority packets being ensured by setting higher maximum occupancy levels for packets of higher priorities, dropping a packet arriving at the buffer if the current occupancy level of the buffer for the priority of the packet concerned is higher than the permitted maximum occupancy level for that priority.

4. A soft handover system for a mobile access network using packet transmission and having a mobile host with a plurality of active soft handover connections (macrodiversity) to the access network, the soft handover connections having respective independent radio paths from the mobile host (MS) to a splitting/recombination node (RNC) of the mobile access network, each radio path having a radio channel to/from the mobile host (MS) to the node (BS, Node B) of the mobile access network which is in radio signal communication with the mobile host (MS), and a transport channel to/from that radio signal communication node (BS, Node B) to the splitting/recombination node (RNC), the system comprising:-

quality measurement apparatus which is operable to measure the quality of the radio channel at the radio communication signal node (BS, Node B) concerned, for each of the soft handover radio paths;

a decision unit which is operable to make a transport priority decision in the splitting/recombination node (RNC), the decision giving higher priorities to the

soft handover connections having higher quality radio
channels;

classification equipment which is operable to effect
packet classification at the or each sending node, a
sending node being a radio signal communication node
(BS, Node B) in the uplink direction, and the sending
node being the splitting/recombination node (RNC) in
the downlink direction, each packet to be sent being
marked according to the assigned priority level of the
radio channel via which it has travelled or is to
travel; and

a buffer manager which is operable to effect buffer
management in all nodes handling the marked packets,
in accordance with the assigned priority levels of the
packets, to ensure that a lower packet loss ratio is
experienced by higher priority packets.

5. A system as claimed in claim 4, wherein the buffer
manager is so operable as to ensure that packets
belonging to the same soft handover connection are not
reordered even if they have different priorities.

6. A system as claimed in claim 4 or 5, wherein the
buffer manager employs, in a sending node concerned,
a single First-In First-Out buffer for all packets,
all priorities and all connections of a given service
type, priority handling of higher priority packets
being ensured by setting higher maximum occupancy
levels for packets of higher priorities, dropping a
packet arriving at the buffer if the current occupancy
level of the buffer for the priority of the packet
concerned is higher than the permitted maximum
occupancy level for that priority.
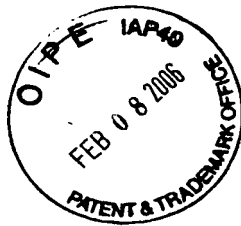
Application No: GB 0024177.8
Claims searched: All

Examiner: Gareth Griffiths
Date of search: 19 June 2001

Patents Act 1977
Search Report under Section 17

Databases searched:

UK Patent Office collections, including GB, EP, WO & US patent specifications, in:

UK Cl (Ed.S): H4L (LDDSX, LDGP, LRPMS, LRRMS)

Int Cl (Ed.7): H04L 12/56, H04Q 7/22, 7/24, 7/38

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